

8. Interpret the statement "Rounding is preferred than truncation in realizing the digital filter".
9. Name the functional units in a Digital Signal Processor and list their features.
10. Illustrate circular buffering in DSPs with an example.

PART B — (5 × 13 = 65 marks)

11. (a) (i) Explain any four properties of DFT. (7)
- (ii) Find the 8-point DFT of the sequence $x[n] = \{0,1,2,3,4,5,6,7\}$ using Decimation in Frequency FFT algorithm. (6)

Or

- (b) (i) Explain the Radix-2 Decimation in Time FFT algorithm. (7)
 - (ii) Find the linear convolution of finite duration sequence $h[n] = [1,2]$ and $x[n] = [1,2,-1,2,3,-2,-3,-1,1,2,1]$ using overlap save method. (6)
12. (a) (i) Utilize Bilinear transformation to design a digital Chebyshev filter for the following specifications (7)

$$0.707 \leq |H(e^{j\omega})| \leq 1 \quad 0 \leq \omega \leq 0.2\pi$$

$$|H(e^{j\omega})| \leq 0.1 \quad 0.5\pi \leq \omega \leq \pi$$

Assume $T = 1$ sec.

- (ii) Make use of direct form I and direct form II structures to realize the system.
- $$y[n] = -0.1y[n-1] + 0.2y[n-2] + 3x[n] + 3.6x[n-1] + 0.6x[n-2] \quad (6)$$

Or

- (b) (i) Describe the steps to design a digital filter using the Impulse Invariance Method. (7)
- (ii) Using impulse invariance method, determine $H(z)$ for the analog transfer function $H(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$. Assume $T = 1$ sec. (6)

13. (a) Use frequency sampling method to determine the impulse response $h(n)$ of a filter with $N = 7$. The desired response is given by (13)

$$H_d(\omega) = \begin{cases} e^{-j3\omega} & 0 \leq |\omega| \leq \frac{\pi}{2} \\ 0 & \frac{\pi}{2} \leq |\omega| \leq \pi \end{cases}$$

Find the transfer function of the filter and model it using minimum number of multipliers.

Or

- (b) (i) Explain the steps in the design of linear phase FIR filters using Fourier series method. (7)

- (ii) Model the transfer function of FIR filter $H(z) = 1 + \frac{3}{4}z^{-1} + \frac{17}{8}z^{-2} + \frac{3}{4}z^{-3} + z^{-4}$ using direct form and cascade form realization. (6)

14. (a) Interpret the effect of Quantization errors in computation of DFT and FFT algorithms. (13)

Or

- (b) An LTI system is characterized by the difference equation $y(n) = 0.95y(n-1) + x(n)$. Infer the limit cycle behavior and determine the dead band of the system when (13)

$$x(n) = \begin{cases} 0.875 & \text{for } n = 0 \\ = 0 & \text{otherwise} \end{cases}$$

Assume that the product is quantized to 4 bits (excluding sign bit) by rounding.

15. (a) With flow diagram, describe the data path and MAC unit in a DSP Processor. (13)

Or

- (b) Classify the addressing modes used in digital signal processors and explain them with examples. (13)

PART C — (1 × 15 = 15 marks)

16. (a) Using Hamming window, design an ideal High pass filter for the frequency response (15)

$$H_d(e^{j\omega}) = 1 \text{ for } \frac{\pi}{4} \leq |\omega| \leq \pi$$

$$= 0 \text{ for } |\omega| \leq \frac{\pi}{4}$$

Compute the values of $n(n)$ for $N = 11$ and determine its transfer function $H(z)$.

Or

- (b) Design a digital Butterworth filter to satisfy the following constraints using bilinear transformation. Assume $T = 1s$. (15)

$$0.9 \leq |H(e^{j\omega})| \leq 1 \text{ for } 0 \leq \omega \leq \pi/2$$

$$|H(e^{j\omega})| \leq 0.2 \text{ for } 3\pi/4 \leq \omega \leq \pi$$

Analyze the poles of the transfer function obtained and assess the stability of the filter.